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2654 #
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PATENT

IN THE UNITED STATES PATENT AND TRADEMARK OFFICE

In re application of: Gersho et al.

Application No. 09/401,068

Filed: September 22, 1999

Confirmation No. 4146

For: LPC-HARMONIC VOCODER WITH
SUPERFRAME STRUCTURE

Examiner: Angela A. Armstrong

Art Unit: 2654

Attorney Reference No. 3382-55844-01

CERTIFICATE OF MAILING

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Copies of United States patents and United States published patent applications do not have to be provided to the Patent Office (37 C.F.R. 1.98(a)(2)(ii)). Copies of unpublished U.S. applications do not have to be provided, as long as the application is available on PAIR, as this requirement of 37 C.F.R. § 1.98(a)(2)(iii) has been waived by the United States Patent and

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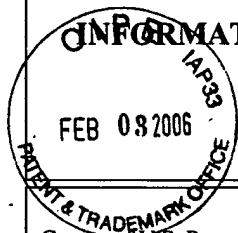
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		Japan	JP 1013200 (abstract)	Jan-89	Boisukurafuto Inc
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		WIPO	WO 98 27543	Jun-98	Interval Research Corporation

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